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For: DIRECTION CONVERSION RECEIVER AND RECEIVING METHOD

CLAIM FOR PRIORITY UNDER 35 USC § 119

Commissioner for Patents
P.O. Box 1450
Alexandria, VA 22313-1450

January 14, 2004

Sir:

The benefit of the filing date of the following prior foreign application filed in the following foreign country is hereby requested for the above-identified patent application and the priority provided in 35 U.S.C. §119 is hereby claimed:

Finnish Patent Application No. 20031194 filed on August 28, 2003 in Finland

In support of this claim, certified copy(ies) of said original foreign application(s) is/are filed herewith.

It is requested that the file of this application be marked to indicate that the requirements of 35 U.S.C. §119 have been fulfilled and that the Patent and Trademark Office kindly acknowledge receipt of these/this document(s).

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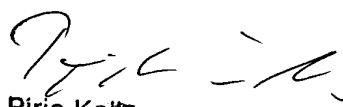
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Keksinnön nimitys
Title of invention

"Direct conversion receiver and receiving method"
(Suoramuuntovastaanotin ja vastaanottomenetelmä)

Täten todistetaan, että oheiset asiakirjat ovat tarkkoja jäljennöksiä Patentti- ja rekisterihallitukselle alkuaan annetuista selityksestä, patenttivaatimuksista, tiivistelmästä ja piirustuksista.

This is to certify that the annexed documents are true copies of the description, claims, abstract and drawings originally filed with the Finnish Patent Office.


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Direct conversion receiver and receiving method

Field

The invention relates to a direct conversion receiver and a reception method in direct conversion receivers. The invention relates particularly to removal of IQ phase imbalance in direct conversion receivers.

Background

The use of digital wireless communication systems has recently been increasing. Systems of many different types have been introduced. For example, systems like Wireless LANs (Local Area Networks), UMTS (Universal Mobile Telecommunications System) and GSM (Global System for Mobile Communication) are gaining more attention and users are given more alternatives in wireless communication. To get customers interested in new services it is essential that the equipment needed in order to use the services should be priced correctly. Receivers with low cost and low power consumption are thus needed.

A solution to provide affordable receivers with low power consumption is to use a direct conversion analog front-end architecture in the receivers. In the direct conversion solution, a received RF signal is mixed directly into the base band and thereafter analog-to-digital converted. For the mixing process, two signals, a sine and a cosine signal, have to be provided. Because of technical reasons the precise orthogonality of both sinusoidal signals cannot be guaranteed; therefore an angle $\phi \neq 90^\circ$ is measurable between the sine and cosine functions. This phenomenon is commonly called IQ phase imbalance.

Analogue base band components, such as low-pass filters and base-band amplifiers are installed twice: one component for the I-branch and one component for the Q-branch. Because of manufacturing tolerances, or temperature influences, each component of a certain functional type may behave slightly differently compared with its counterpart on the other branch. The conjunction of frequency dependent base band devices with the constant IQ phase imbalance imperfections result in frequency selective IQ phase imbalance inaccuracies.

Because of the IQ imbalance, so-called image signals are created at image frequencies of the actual signals, causing IQ-crosstalk. Figures 1A and 1B show an example of the reception of a multi-carrier signal under the effect of IQ imbalance. As shown in Figure 1A, the received signal contains six

carriers 100 to 114, out of which carriers 100 to 104 form the lower sideband and carriers 110 to 114 above the local oscillator (LO) frequency form the upper sideband. Figure 1B shows how the signal spectrum of Figure 1A may look after frequency conversion to baseband when sufficient image rejection is not performed. In the situation shown by Figure 1B, image signals 110A-114A overlap the converted signals 100A-104A.

For example, in case of two carriers with significant power difference in the received signal, an image of the first component appears in the second component and vice versa. The strengths of the image signals depend on the degree of the imbalance and on the strengths of the signals. Weak image signals are usually not a problem when the signal components have about the same magnitude. However, problems may arise when one signal is strong and the other signal is weak. If the image of the strong signal is stronger than the weak signal, or even comparable to it, the reception of the weak signal is degraded or even entirely prevented. In the example of Figure 1B, image signal 110A of signal 110 is stronger than the weak signal 104A which may prevent reception of signal 104A.

Brief description of the invention

An object of the invention is to provide an improved receiving method and a receiver so that IQ imbalance is reduced thus enhancing reception of weak signal components. According to the invention, there is provided a receiving method in a direct conversion receiver, the method comprising steps of receiving a signal comprising multiple components at different receiving frequencies belonging to a frequency band, mixing each of the received signal components into a corresponding base band signal comprising I- and Q branches, converting the analog base band signal into a digital signal. The method also comprises measuring power levels of the signal components in the digital signal in pairs, where one component in the pair belongs to an upper sideband of the frequency band and one component in the pair belongs to a lower sideband of the frequency band, estimating, when either the upper sideband component or the lower sideband component dominates in power over the other component in the pair, a frequency-independent phase imbalance, a frequency-dependent phase imbalance and a gain imbalance, and compensating the estimated frequency-independent phase imbalance, the frequency-dependent phase imbalance and the gain imbalance to at least one of the I- and Q-branch signals.

The invention also relates to a direct conversion receiver, comprising means for receiving a signal comprising multiple components at different receiving frequencies belonging to a frequency band, means for mixing each of the received signal components into a corresponding base band signal comprising I- and Q branches, means for converting the analog base band signal into a digital signal. The receiver comprises means for measuring power levels of the signal components in the digital signal in pairs, where one component in the pair belongs to an upper sideband of the frequency band and one component in the pair belongs to a lower sideband of the frequency band, means for estimating, when either the upper sideband component or the lower sideband component dominates in power over the component in the pair, a frequency-independent phase imbalance, a frequency-dependent phase imbalance and a gain imbalance, and means for compensating the estimated frequency-independent phase imbalance, the frequency-dependent phase imbalance and the gain imbalance to at least one of the I- and Q-branch signals.

The object of the invention is also to provide a direct conversion receiver, comprising a receiver to receive a signal comprising multiple components at different receiving frequencies belonging to a frequency band, a mixer to mix each of the received signal components into a corresponding base band signal comprising I- and Q branches, an analog-to-digital converter to convert the analog base band signal into a digital signal. The receiver comprises a measuring unit to measure power levels of the signal components in the digital signal in pairs, where one component in the pair belongs to an upper sideband of the frequency band and one component in the pair belongs to a lower sideband of the frequency band, an estimator to estimate, when either the upper sideband component or the lower sideband component dominates in power over the component in the pair, a frequency-independent phase imbalance, a frequency-dependent phase imbalance and a gain imbalance and a compensator to compensate the estimated frequency-independent phase imbalance, the frequency-dependent phase imbalance and the gain imbalance to at least one of the I- and Q-branch signals.

Preferred embodiments of the invention are described in the dependent claims.

The invention thus relates to direct-conversion receivers, where a received RF signal is directly mixed into a base band signal without analog

down-conversion to intermediate frequency (IF). The analog base band signal is thereafter analogue-to-digital converted. In the invention, the IQ-phase imbalance is determined and the correction for the imbalance is performed into a digital base band signal.

5 A multicarrier signal can be divided into pairs occupying the same (positive and negative) baseband frequency bands as shown in Figure 1A. For example, in situation of Figure 1A, there are 6 carriers and correspondingly there are then 3 carrier pairs. According to one aspect of the invention, power levels of the received signal components are measured and the estimation for the IQ imbalance are performed pairwise when and only when either the upper or the lower sideband signal component of the carrier pair dominates over the other component. Once estimation has been carried out, compensation can but does not need to be carried out when signals are equal.

10 Estimation for each pair can be carried out independently, i.e., estimation can be done for pair #1 if one of two carriers in it dominates, and likewise for pairs #2 and #3. In one embodiment, the I- and Q-signals are transformed into frequency domain, and the estimation of IQ imbalance is based on phase of the cross-spectrum of the transformed signals.

15 The inventive algorithm and apparatus can be implemented so that one of the signal branches, I and Q, is kept constant, and the phase imbalance is estimated in relation to the branch that is kept constant. In one embodiment, component-specific frequency-dependent phase imbalance estimates are formed and the frequency-independent phase imbalance is estimated as average of the component-specific estimates. In one embodiment, the frequency-dependent phase imbalance is estimated as half of the difference between the component-specific, frequency-dependent phase imbalance estimates.

20 According to one aspect of the invention, the gain imbalance and the frequency-dependent phase imbalance are compensated for by digital filtering, and a frequency-independent phase error is removed by subtracting one of the suitably weighted signal branches I and Q from the other branch signal.

25 The invention can in an advantageous manner compensate for the frequency-dependent IQ imbalance thus reducing disadvantages of IQ crosstalk.

List of drawings

In the following, the invention will be described in greater detail with reference to the preferred embodiments and the accompanying drawings, in which

- 5 Figure 1A illustrates reception of a multicarrier signal,
- Figure 1B shows the effect of the image signals in the frequency domain,
- Figure 2A shows one embodiment of the method according to the invention,
- 10 Figure 2B specifies the method disclosed in Figure 2A, and
- Figure 3 illustrates an example of a direct conversion receiver according to an embodiment of the invention.

Description of embodiments

- 15 The embodiments of the invention can be applied in any data transmission system employing direct conversion receivers. Examples of such systems include Wireless LANs, UMTS and GSM. A direct conversion receiver is a receiver where a received radio frequency (RF) signal is converted directly to a base band frequency without any intermediate frequency (IF) conversion in between.

- 20 Figures 2A and 2B disclose one embodiment of the method according to the invention. In Figure 2A step 200, a signal having multiple frequency components is received. The signal can thus originate from a single source or can contain multiple carriers from independent sources. The IQ imbalance problem will here be explained using dual-carrier scenario as an example. However, both the problem and the compensation algorithm can be
- 25 straightforwardly generalized to scenarios with more than two carriers.

- Let $v_{I1}(t)$ and $v_{Q1}(t)$ denote the baseband in-phase and quadrature-phase components of the signal 1, respectively, and let $v_{I2}(t)$ and $v_{Q2}(t)$ be the respective components of the signal 2. The frequency bands of the signals are
- 30 located symmetrically around the local oscillator (LO) frequency f_{LO} . The carrier 1 has thus the frequency $f_{LO}+f_0$ and the carrier 2 has the frequency $f_{LO}-f_0$.

- In the receiving step 200, the dual-carrier signal entering the RF port of the receiver further goes through a bandpass filter and a low-noise amplifier. Then the signal is down-converted, that is, mixed into base band signal
- 35 in a quadrature demodulator according to step 202. The resulting I- and Q-

branch signals can be low-pass filtered, amplified, and converted into a digital signal according to step 204. If the receiver is ideal, the signals $v_I(t)$ and $v_Q(t)$ in the output of the ADC's (Analog-to-Digital Converter) have the form shown by formula (1):

$$\begin{aligned} i(t) &= [v_{I1}(t) + v_{I2}(t)] \cos(2\pi f_0 t) - [v_{Q1}(t) - v_{Q2}(t)] \sin(2\pi f_0 t) \\ q(t) &= [v_{I1}(t) - v_{I2}(t)] \sin(2\pi f_0 t) + [v_{Q1}(t) + v_{Q2}(t)] \cos(2\pi f_0 t) \end{aligned} \quad (1)$$

For clarity of the presentation, phase offset between the RF signal and the LO is ignored as irrelevant. To resolve the signals 1 and 2, the digital signals $i(t)$ and $q(t)$ are subjected to another quadrature down-conversion, in which they are mixed with signals $\cos(2\pi f_0 t)$ and $\sin(2\pi f_0 t)$. Lowpass-filtering the mixing products and combining them appropriately yields the desired signals $v_{I1}(t)$, $v_{Q1}(t)$, $v_{I2}(t)$ and $v_{Q2}(t)$.

Now consider that the receiver has some IQ imbalance. Part of the imbalance arises from the I- and Q-branch LO signals differing from perfect phase quadrature. This imbalance is in the present discussion called frequency-independent phase imbalance. In addition to the frequency-independent phase imbalance, there may be frequency-dependent deviations in the gain and phase responses of the I- and Q-branch mixers, low-pass filters, and other analog components. Without loss of generality, one may take the Q-branch of the receiver as reference and assume that all nonideality occurs in the I-branch only.

In step 206, powers of signal components are measured and step 208 means that powers of the signal components on the opposite sides of the LO frequency are pair-wise compared to each other. When one of the signal components in the pair dominates in power over the other signal component in the pair, the method proceeds over to estimating of phase and gain imbalance according to step 210. The estimation provides the correcting step 212 with correction factors to correct the phase and gain imbalance.

Referring to Figure 2B, the invention is now disclosed by specifying the estimating and correcting steps of Figure 2A. For analysis purposes, it is more convenient to express the signals in frequency domain, the transformation from time to frequency domain being performed in step 210A. The transformation can be performed e.g. with Fast Fourier transform or discrete Fourier transform.

Let $I(f)$ and $Q(f)$ denote the Fourier transforms of the ideal signals $i(t)$ and $q(t)$, and let $X(f)$ and $Y(f)$, shown by formulas (2), denote the corresponding signals in the presence of IQ imbalance

$$\begin{aligned} X(f) &= \Delta g(f) \exp(j\phi(f)) \{\cos(\delta) I(f) + \sin(\delta) Q(f)\}, \\ Y(f) &= Q(f). \end{aligned} \quad (2)$$

Here, $\Delta g(f)$ and $\phi(f)$ denote, respectively, gain and phase imbalance as functions of frequency, and δ is phase error between the I- and Q-branch LO signals. When the signals $X(f)$ and $Y(f)$ are subjected to digital down-conversion, the outcome is not only signals 1 and 2. In addition, an image of the signal 1 appears in signal 2 and vice versa. The strengths of the image signals depend on the degree of the imbalance and on the strengths of the signals.

Note that steps 210A-210D are not limited to the order of steps but on the contrary, these steps can be performed in any order. In the following, the above mentioned dual-carrier scenario is still used as an example. However, the method can be straightforwardly generalized to situations with more than two carriers.

Referring to the IQ imbalance given in Eq. (2), it can be seen that if the IQ imbalance parameters δ , $\Delta g(f)$ and $\Phi(f)$ are known, estimates $I'(f)$ and $Q'(f)$ for the desired signals $I(f)$ and $Q(f)$, respectively, can be computed from the measured signals $X(f)$ and $Y(f)$ as shown by formulas (3)

$$\begin{aligned} I'(f) &= \exp(-j\Phi(f)) / (\cos(\delta) \Delta g(f)) X(f) - \tan(\delta) Y(f), \\ Q'(f) &= Y(f). \end{aligned} \quad (3)$$

In method step 210B, frequency-dependent phase imbalance $\Phi(f)$ is estimated. The estimation is done from the complex-valued frequency-domain signals $X(f)$ and $Y(f)$. Consider a dual-carrier signal of the form (1). The base-band signal components $v_{11}(t)$, $v_{Q1}(t)$, $v_{12}(t)$ and $v_{Q2}(t)$ forming the dual-carrier signal are assumed to be noise-like in the sense that their Fourier-transforms, $V_{11}(f)$, $V_{Q1}(f)$, $V_{12}(f)$ and $V_{Q2}(f)$, respectively, satisfy formulas (4)

$$\begin{aligned} \langle V_{1a}(f) V_{1b}(f) \rangle &= 0, & \text{where } a = 1, 2 \text{ and } b = 1, 2, \\ \langle V_{1a}(f) V_{Qb}(f) \rangle &= 0, & \text{where } a = 1, 2 \text{ and } b = 1, 2, \\ \langle V_{Qa}(f) V_{Qb}(f) \rangle &= 0, & \text{where } a = 1, 2 \text{ and } b = 1, 2. \end{aligned} \quad (4)$$

Here, notation $\langle Z \rangle$ denotes the expectation value of the quantity Z , evaluated in practice by averaging the values of Z over several sets of data. Moreover, assume that the signals 1 and 2 are independent, in the sense that they satisfy formula (5)

$$\langle V_{a1}(f) V_{b2}^*(f) \rangle = 0, \text{ where } a = I, Q \text{ and } b = I, Q. \quad (5)$$

Here, asterisk $*$ denotes complex conjugation. The in-phase and quadrature-phase components $v_{I1}(t)$ and $v_{Q1}(t)$ of the signal 1 are not required to be uncorrelated, neither are those of the signal 2. With some algebraic manipulation, it can be shown that formula (6, continues on two lines) applies

$$\begin{aligned} \langle X(f) Y^*(f) \rangle &= j^{1/4} \Delta g(f) \exp(j\Phi(f)) \times \\ &(\exp(-j\delta)(S_1(f_0 + f) + S_2(f - f_0)) - \exp(j\delta)(S_1(f - f_0) + S_2(f_0 + f))). \end{aligned} \quad (6)$$

Formula (7) shows $S_1(f)$ and $S_2(f)$ respectively denoting the spectra of the signals 1 and 2:

$$\begin{aligned} S_a(f) &= \langle (V_{Ia}(f) + jV_{Qa}(f)) (V_{Ia}(f) + jV_{Qa}(f))^* \rangle \\ \text{where } a &= 1, 2. \end{aligned} \quad (7)$$

Assuming that spectra are sufficiently band-limited, following formula (8) applies:

$$\langle X(f_0 + f) Y^*(f_0 + f) \rangle \cong j^{1/4} \Delta g(f_0 + f) \exp(j\delta(f_0 + f)) (\exp(-j\delta) S_2(f) - \exp(j\delta) S_1(f)). \quad (8)$$

Consequently, the signal 1 with RF carrier at frequency $f_{LO} + f_0$, experiences an effective phase imbalance $\phi_1(f)$, whereas the signal 2, with RF carrier frequency $f_{LO} - f_0$, experiences an effective phase imbalance $\phi_2(f)$, the phase imbalances shown by equations (9)

$$\begin{aligned} \phi_1(f) &= \delta + \Phi(f), \\ \phi_2(f) &= \delta - \Phi(f). \end{aligned} \quad (9)$$

Hence, estimation of the phase imbalance from the phase of the averaged quantity $\langle X(f)Y^*(f) \rangle$ yields considerable advantages. Because of averaging, the phase of cross-products of signals satisfying formulas (4) and (5) oscillates randomly and rapidly vanishes, leaving only contributions proportional to the spectrum of the signals to remain, as shown in formula (8). Consequently, correlation between the I- and Q-components of signal 1 or those of signal 2 only affects the spectrum of those signals and does not influence the phases. Moreover, since the phase of the terms proportional to the spectrum of signals 1 and 2 are independent of the exact shape of the spectra, the signals themselves do not need to be known and the estimation is blind, i.e., the algorithm does not require pilot signals or decision-aided feedback. Finally, the phase of the quantity $\langle X(f)Y^*(f) \rangle$ directly yields phase imbalance data in Fourier domain. Hence, it is not necessary to use loops in time domain, and the method converges remarkably rapidly. A secondary algorithm is needed to separate the frequency independent and frequency dependent contribution. Several methods to accomplish this separation are included in this invention. These are discussed next.

When both signals 1 and 2 are present, the phase of the averaged quantity $\langle X(f)Y^*(f) \rangle$ depends on the relative strengths of the signals. However, under the condition that either $S_1(f) \gg S_2(f)$ or $S_2(f) \gg S_1(f)$, the phase of the averaged quantity $\langle X(f)Y^*(f) \rangle$ directly yields either the quantity $\phi_1(f)$ or the quantity $\phi_2(f)$. Referring back to Figure 2A and steps 206 and 208 the phase imbalance is estimated from the phase of the averaged quantity $\langle X(f)Y^*(f) \rangle$ only when one of the signals in the signal component pair dominates in receiving power over the other one. In optimal conditions the dominating signal alternates over a reasonable measuring period, i.e., signal 1 dominates part of the time and signal 2 dominates part of the time. In such conditions, estimates are obtained for both quantities $\phi_1(f)$ and $\phi_2(f)$. Referring again to Figure 2B and step 210C, the phase error between I and Q LO-signals δ , that is, the frequency independent phase error, can be estimated according to formula (10) as an average

$$\delta = (\phi_1(f) + \phi_2(f))/2 \quad (10)$$

over the frequency band of interest, and frequency-dependent phase error $\Phi(f)$ of step 210B can be estimated according to formula (11)

$$\Phi(f) = (\varphi_1(f) - \varphi_2(f))/2. \quad (11)$$

Alternatively, for example polynomial fitting methods can be used to
5 extract the imbalances from either quantity $\varphi_1(f)$ or $\varphi_2(f)$ alone.

One approach to have the desired alternation in the relative signal
powers is to use the RF test loop. In unit testing/calibration stage, signals gen-
erated by the RF test loop can be used to achieve initial values for the imbal-
ance. During operation, if there are time periods when only one of the signals
10 is being received, the RF test loop could be used to generate the other signal
for imbalance estimation purposes.

The method can be straightforwardly generalized to cases with
more than two carriers. It is then required that the phase imbalance is esti-
mated only when each frequency band of interest is dominated by only one of
15 the two possible carriers that can occupy that particular frequency band. De-
composition into frequency-dependent and frequency-independent imbalance
contribution is then straightforward.

Estimation of gain imbalance $\Delta g(f)$ as shown by step 210D, can be
estimated with some prior art method utilizing the signals from ADCs to obtain
20 frequency-domain signals $X(f)$ and $Y(f)$ by Fourier transform according to step.
As a result of the Fourier transform, the amplitude levels of each frequency
component are determined. Initially, both I- and Q-branch contain both side-
bands but with different phases. In the average, the I- and Q-branches have
the same power, if both branches are identical. The ratio of the amplitude av-
25 erages at each frequency point of the I- and Q-branches reveals the frequency
dependent gain difference between the I- and Q-branch of the receiver.

Method steps 210B-210D presented above contribute to determin-
ing of, according to step 210E, the desired signals presented by equations (3).
As a practical realization, e.g. a digital equalizer such as a FIR (Finite Impulse
30 Response) filter can be realized so that its frequency response matches the
inverse imbalance $\exp(-j\Phi(f)) / (\cos(\delta) \Delta g(f))$. The accuracy of the correction
depends on the accuracy of the estimated imbalance parameters. Hence, the
algorithm for imbalance estimation is of crucial importance.

As a second correction step, after step 212A when correcting with
35 an equalizer compensating the gain imbalance and frequency-dependent
phas imbalance, a second correction step 212B is introduced, where correc-

tion is done with a subtraction element compensating the frequency-independent phase imbalance. Tunable delay elements can be added to control overall delay between the I- and Q-branches and to alleviate the filter synthesis.

5 An alternative solution, in comparison to the one disclosed above, is, not to use delay elements but, instead, to carry out the correction to either I- or Q-branches according to which branch has a shorter delay.

The inventive idea can also be implemented so that LO phase error is compensated by introducing an opposite phase imbalance into the digital oscillator in the baseband double quadrature stage which follows the compensation stage.

Figure 3 illustrates an example of a receiver according to the invention. The receiver comprises receiving means 300 that comprises an antenna, whereby a radio frequency signal from a transmitter is received. The received signal is band pass filtered in a filter 302 and amplified in a low noise amplifier 304. Then, the signal is mixed directly into the base band frequency in mixers 306A-306B where local oscillator 308 provides the RF signal used in the mixing. After the mixing process, an analog complex signal $s(t) = I(t) + jQ(t)$, having approximately 90 degrees phase difference between the signal branches, is available. I- and Q-components are passed to corresponding analogue low pass filters 310A-310B and a base band amplifiers 312A-312B. The signals are then converted into digital signals in converting means, that is, A/D converters 314A-314B. These receiver components are all known to a man skilled in the art.

In multicarrier systems, IQ phase errors have a significant impact on the signal performance, compared with single carrier radios. It is therefore important to eliminate the IQ phase imbalance in the receiver as early as possible. In the embodiment illustrated by the receiver of Figure 3, an IQ phase imbalance estimation block 316 follows directly after the A/D converters 314A-314B. The signal is then processed by tunable delay elements 318A-318B to control overall delay between the I- and Q-branches and to alleviate the filter synthesis. In the estimation block 316, frequency-dependent phase imbalance $\Phi(f)$, frequency-independent phase imbalance δ and gain imbalance $\Delta g(f)$ are estimated. Figure 3 shows one example to correct the estimated factors of the total IQ imbalance. The receiver contains a compensating means, which comprises a first correcting means 320A such as a FIR filter to compensate for the frequency-dependent phase imbalance $\Phi(f)$ and the gain imbalance $\Delta g(f)$.

The compensating means further contains a subtraction element 320B-320C to subtract the frequency-independent phase imbalance δ to obtain the compensated signals $I'(f)$ and $Q'(f)$ in equations (3).

5 The invention can be realized as a DSP (Digital Signal Processing) algorithm. For example, the invention can be implemented together with the other RF-related DSP functions, such as digital down-conversion, filtering and scaling. For example, ASIC (Application Specific Integrated Circuit), FPGA (Field Programmable Gate Arrays) or processor-based approaches can be used.

10 Even though the invention has been described above with reference to an example according to the accompanying drawings, it is clear that the invention is not restricted thereto but can be modified in several ways within the scope of the appended claims.

Claims

1. A receiving method in a direct conversion receiver, the method comprising:

5 receiving a signal comprising multiple components at different receiving frequencies belonging to a frequency band;

mixing each of the received signal components into a corresponding base band signal comprising I- and Q branches;

converting the analog base band signal into a digital signal;

10 characterized by

measuring power levels of the signal components in the digital signal in pairs, where one component in the pair belongs to an upper sideband of the frequency band and one component in the pair belongs to a lower sideband of the frequency band;

15 estimating, when either the upper sideband component or the lower sideband component dominates in power over the other component in the pair, a frequency-independent phase imbalance, a frequency-dependent phase imbalance and a gain imbalance;

20 compensating the estimated frequency-independent phase imbalance, the frequency-dependent phase imbalance and the gain imbalance to at least one of the I- and Q-branch signals.

2. The method of claim 1, characterized in that when estimating:

25 transforming the I- and Q-signals into frequency domain using a discrete Fourier transform or a fast Fourier transform to provide signals $X(f)$ and $Y(f)$;

estimating the frequency-dependent phase imbalance and the frequency-independent phase imbalance from the phase of the cross-spectrum $(X(f)Y^*(f))$.

30 3. The method of claim 2, characterized by

estimating the frequency-dependent phase imbalance and the frequency-independent phase imbalance from the phase of the averaged cross-spectrum $\langle X(f)Y^*(f) \rangle$.

4. The method of claim 1, characterized in that in estimating:
estimating signal component-specific frequency-dependent phase imbalances when either the upper or the lower sideband signal component present in the pair dominates in power over the other component;

5 estimating the frequency-independent phase imbalance as an average over the component-specific frequency-dependent phase imbalances.

5. The method of claim 1, characterized in that in the estimating step:

10 estimating signal component-specific frequency-dependent phase imbalance factors when either the upper or the lower sideband signal component in the pair dominates in power over the other component;

estimating the frequency-dependent phase imbalance as a half of the difference between the component-specific frequency-dependent phase imbalance factors.

15 6. The method of claim 1, characterized in that in the estimating step:

estimating signal component-specific frequency-dependent phase imbalances when either the upper or the lower sideband signal component of the pair dominates in power over the other component;

20 estimating the frequency-independent phase imbalance from one or several of the component-specific frequency-dependent phase-imbalances by fitting techniques.

7. The method of claim 1, characterized in that in the compensating step:

25 compensating for the frequency-dependent phase imbalance and for the gain imbalance by digital filtering.

8. The method of claim 7, characterized in that in the compensating step:

30 compensating for the frequency-independent phase imbalance by subtracting the frequency independent phase imbalance from the outcome of the digital filtering.

9. A direct conversion receiver, comprising:

means for receiving a signal comprising multiple components at different receiving frequencies belonging to a frequency band;

35 means for mixing each of the received signal components into a corresponding base band signal comprising I- and Q branches;

means for converting the analog base band signal into a digital signal;

characterized in that the receiver comprises:

means for measuring power levels of the signal components in the digital signal in pairs, where one component in the pair belongs to an upper sideband of the frequency band and one component in the pair belongs to a lower sideband of the frequency band;

means for estimating, when either the upper sideband component or the lower sideband component dominates in power over the component in the pair, a frequency-independent phase imbalance, a frequency-dependent phase imbalance and a gain imbalance; and

means for compensating the estimated frequency-independent phase imbalance, the frequency-dependent phase imbalance and the gain imbalance to at least one of the I- and Q-branch signals.

10. The direct conversion receiver of claim 9, characterized in that the estimating means is configured to:

transform the I- and Q-signals into frequency domain using discrete Fourier transform or fast Fourier transform to provide signals $X(f)$ and $Y(f)$;

estimate the frequency-dependent phase imbalance and the frequency-independent phase imbalance from the phase of the cross-spectrum $(X(f)Y^*(f))$.

11. The direct conversion receiver of claim 10, characterized by

estimating the frequency-dependent phase imbalance and the frequency-independent phase imbalance from the phase of the averaged cross-spectrum $\langle X(f)Y^*(f) \rangle$.

12. The direct conversion receiver of claim 9, characterized in that the estimating means is configured to:

estimate signal component-specific frequency-dependent phase imbalances when either upper- or lower sideband signal component present in the pair dominates in power over the other component;

estimate the frequency-independent phase imbalance as an average over the component-specific frequency-dependent phase imbalances.

13. The direct conversion receiver of claim 9, characterized in that the estimating means is configured to:

estimate signal component-specific frequency-dependent phase imbalance factors when either the upper- or the lower sideband signal component in the pair dominates in power over the other component;

5 estimate the frequency-dependent phase imbalance as a half of the difference between the component-specific frequency-dependent phase imbalance factors.

14. The direct conversion receiver of claim 9, characterized in that the estimating means is configured to:

10 estimate signal component-specific frequency-dependent phase imbalances when either the upper- or the lower sideband signal component of the pair dominates in power over the other component;

estimate the frequency-independent phase imbalance from one or several of the component-specific frequency-dependent phase-imbalance by fitting techniques.

15 15. The direct conversion receiver of claim 9, characterized in that the compensating means is configured to:

compensate for the frequency-dependent phase imbalance and for the gain imbalance by digital filtering.

20 16. The direct conversion receiver of claim 15, characterized in that the compensating means is configured to:

compensate for the frequency-independent phase imbalance by subtracting the frequency independent phase imbalance from the outcome of the digital filtering.

25 17. A direct conversion receiver, comprising:

a receiver to receive a signal comprising multiple components at different receiving frequencies belonging to a frequency band;

a mixer to mix each of the received signal components into a corresponding base band signal comprising I- and Q branches;

30 an analog-to-digital converter to convert the analog base band signal into a digital signal;

characterized in that the receiver comprises:

35 a measuring unit to measure power levels of the signal components in the digital signal in pairs, where one component in the pair belongs to an upper sideband of the frequency band and one component in the pair belongs to a lower sideband of the frequency band;

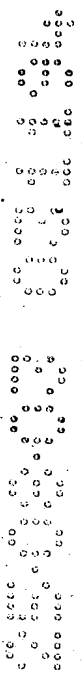
an estimator to estimate, when either the upper sideband component or the lower sideband component dominates in power over the component in the pair, a frequency-independent phase imbalance, a frequency-dependent phase imbalance and a gain imbalance; and

- 5 a compensator to compensate the estimated frequency-independent phase imbalance, the frequency-dependent phase imbalance and the gain imbalance to at least one of the I- and Q-branch signals.

(57) Abstract

A direct conversion receiver, comprising means for receiving a signal comprising multiple components at different receiving frequencies belonging to a frequency band, means for mixing the signal components into a base band signal comprising I- and Q branches, means for converting the analog base band signal into a digital signal. The receiver comprises means for measuring power levels of the signal components in pairs, where one component in the pair belongs to an upper sideband of the frequency band and one component in the pair belongs to a lower sideband of the frequency band, means for estimating, when either the upper sideband component or the lower sideband component dominates in power over the component in the pair, a frequency-independent phase imbalance, a frequency-dependent phase imbalance and a gain imbalance, and means for compensating the estimated imbalances to at least one of the I- and Q-branch signals.

(Fig. 3)



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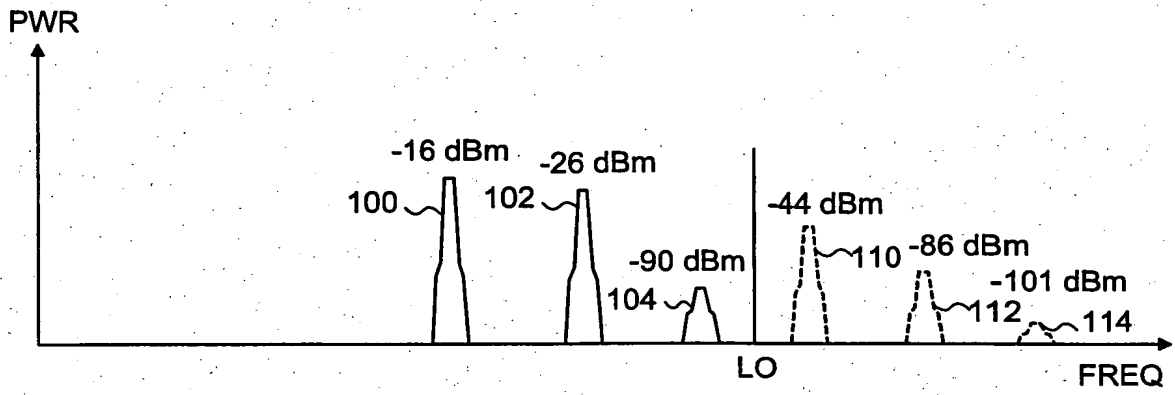


Fig. 1A

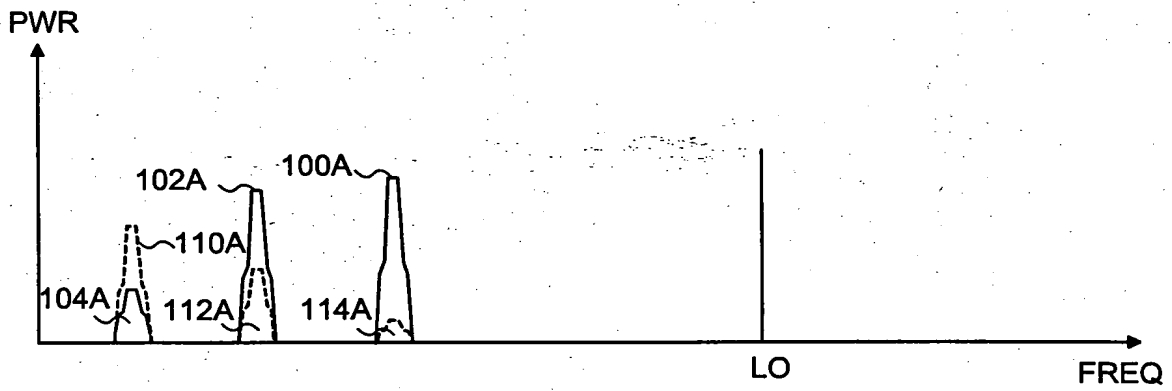


Fig. 1B

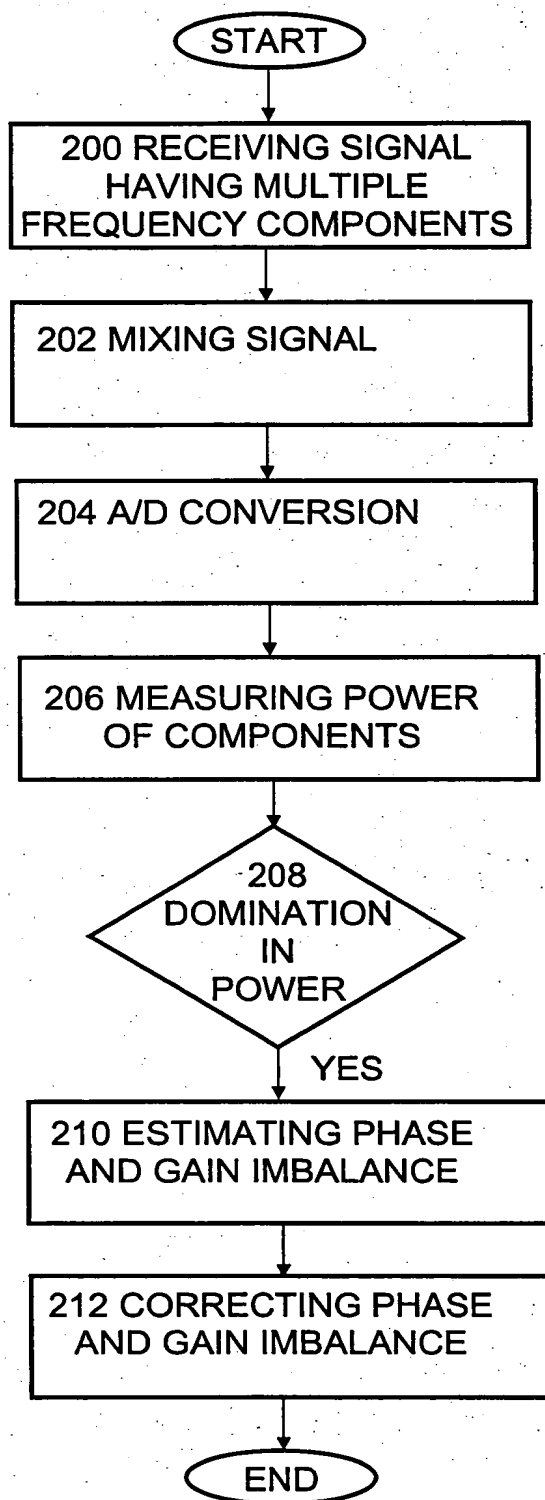


Fig. 2A

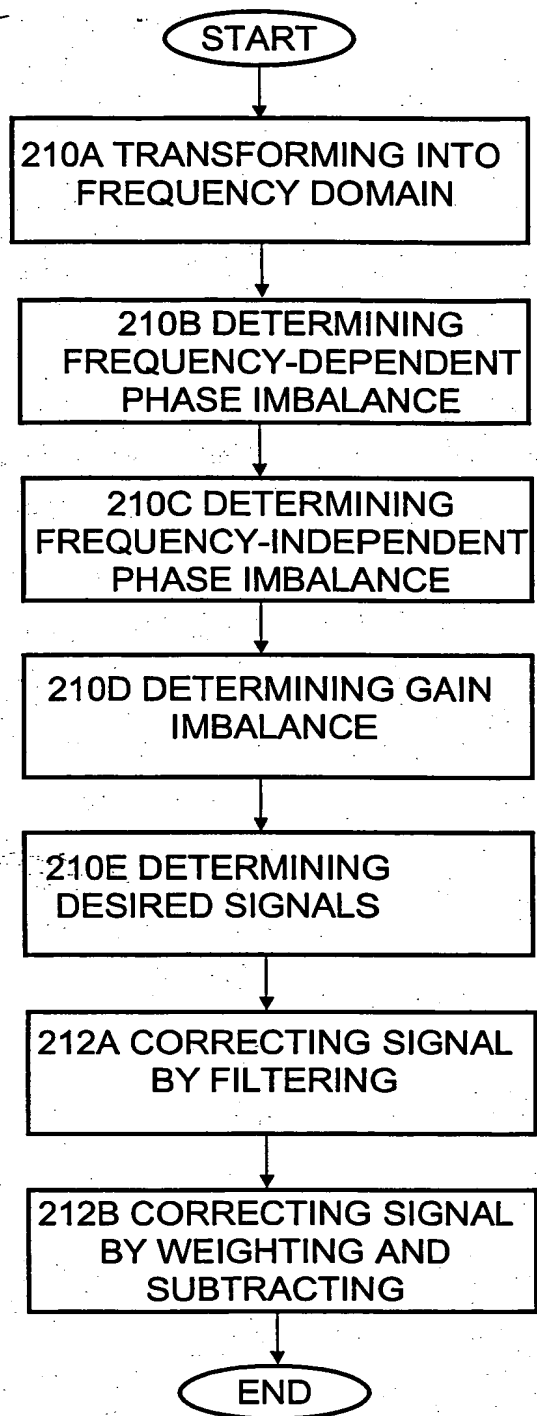


Fig. 2B

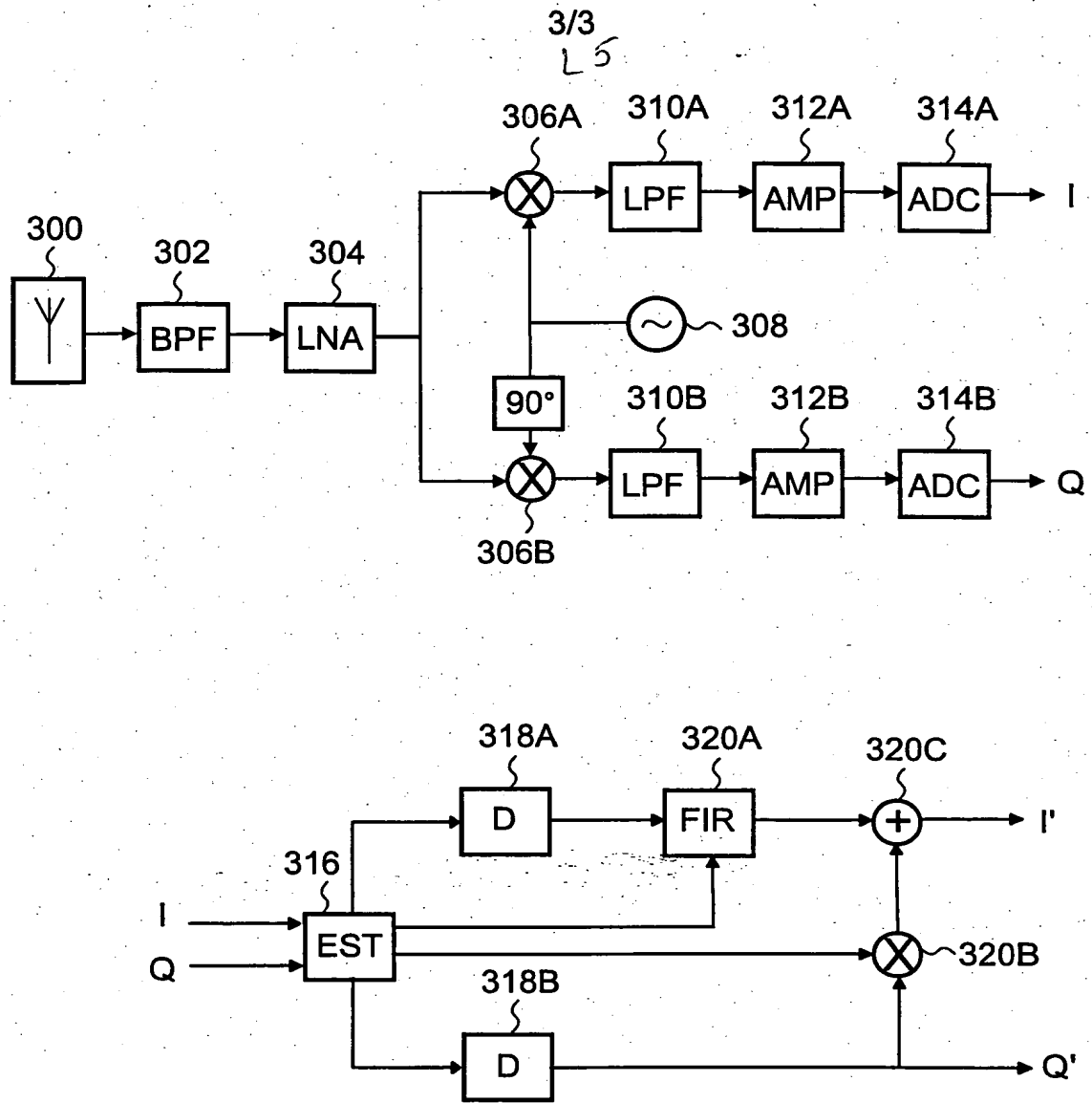


Fig. 3